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**This is a U.S. Patent Application for:**

**Title:**           **APPARATUS AND METHOD FOR DELIVERY OF RINGING  
AND VOICE CALLS THROUGH A WORKSTATION**

**Inventor #1:** Peter Kozdon  
**Address:**    2252 Pyle Court, Santa Clara, CA 95051  
**Citizenship:** United Kingdom

**Inventor #2:** Mark Skrzynski  
**Address:**    1800 42nd Avenue, Capitola, CA 95010  
**Citizenship:** U.S.A.

## APPARATUS AND METHOD FOR DELIVERY OF RINGING AND VOICE CALLS THROUGH A WORKSTATION

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### BACKGROUND OF THE INVENTION

10 Data processing devices such as workstations or personal computers are or can be equipped with an audio card to provide audio input and output functions. Modern multi-media systems use such audio cards to support telephony services using a modem or a network connection rather than the traditional telephone network. Telephony functions using a modem can be either regular "analog"-telephone functions provided by a so-called voice modem, or telephony over data network using a modem or a network interface card to couple to data-networks, such as the Internet.

15 Today's multi-media workstations or personal computers (PC) support a wide variety of different complementary products to enhance these functions, such as loudspeakers, head-sets with microphones, etc. Nevertheless, in many environments, such as offices, it is often required to use a speaker for alerting (ringing the phone) and after connecting either a speaker phone mode for convenience or a handset to provide privacy of a phone call. As usual multi-media system providing telephone functions only use standard audio cards with a single (common) audio output, the user can either use speakers or a headset in the telephone environment. To select between either a headset and a loudspeaker an external switch-box is required. The user must remember to switch back to the speaker mode after 20 the end of the call, to ensure that he can hear the ringing of the next call, depending on the relative sensitivities of the speaker and headset, a change of the volume setting may also be required.

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### SUMMARY OF THE INVENTION

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Therefore, it is an object of the present invention to provide an improved audio input and output means for a data processing system with the capability of addressing either or both a headset or a loudspeaker. This

object is achieved by a data processing system comprising an audio processing means receiving data within said data processing system for converting the data into an analog stereo signal with a first and a second analog channel; an audio output means receiving the analog stereo signal and providing a first and second output signal wherein the first output signal is provided for a loudspeaker and the output second signal is provided for a headset; the audio processing means upon a control signal either provide an audio signal on the first or on the second channel. This method still uses the usual audio soundcards found in most workstations or PC's.

Furthermore, a method provides the steps of: receiving a digital signal representing an audio signal; receiving a control signal; processing the digital signal to generate a two channel signal and upon the control signal providing the signal for either or both a first or a second audio channel, wherein the first signal is provided for a loudspeaker and the second signal is provided for a headset; converting said first and second signals into analog signals.

Upon receiving a ringing signal which can be presented to both the speaker and the headset or only to the speaker, the method allows the user to select either the headset or the loudspeaker as an output medium. The control signal can be generated by a keyboard, a mouse, a trackball or any other input device.

The method allows for independent setting of the output level (volume) for both the speaker and headset, there is no danger of the user being subjected to excessive sound level, when switching between the headset and speaker, or should the connection be left in the wrong state after the completion of a call (loud ringing through headset - or missing the call because the volume is too low).

#### BRIEF DESCRIPTION OF THE DRAWINGS

Fig. 1 shows a multi-media system according to the present invention,;

Fig. 2 shows details of a multi-media system according to Fig. 1;

Fig. 3 shows details of the audio system according to a second embodiment;

Fig. 4 shows details of the audio system according to a third embodiment; and

Fig. 5 shows a Alert-Box which can be displayed on a multi-media system according to the present invention.

#### DESCRIPTION OF THE PREFERRED EMBODIMENT

Fig. 1 shows a network 1, such as the Internet, coupled with a variety of I/O devices. For example, a server 2 is coupled with a private branch exchange 3 (PBX) which can connect to a plurality of workstations 5 or telephone sets 4. Also, an internet phone 6 can be directly coupled with the internet. More workstations 7, 12 can be coupled with the internet through a server 11 with the network 1. Each workstation 7 may be a multi-media personal computer 7 with a respective sound card (not shown), speakers 8, a headset 10, and a microphone 13.

If a telephone call is made from one of the telephone sets 4, 6 or workstations 5, 12 to workstation 7 using a packetized voice data stream, a ringing signal will be sent to workstation 7. According to the present invention, the ringing will be output to the speaker 8. In addition, the ring may be also applied to the headset. Together with the ringing a message may pop up on the screen of the workstation 7 asking the user to choose between a speakerphone function or a headset function. Instead, the user could also activate a predefined keyboard action, mouse click, etc. This could be known to the user and done without the screen pop-up menus. In case the user decides to use the speakerphone function the headset will be disabled and the incoming audio data will be directed to the loudspeaker 8. In case the headset function is chosen, the loudspeaker 8 will be disabled and audio signals will be forwarded to the headset 10. In both scenarios microphone 13 may be used as the input medium. If there are multiple audio inputs on the soundcard, it would be possible to use separate mic's for headset, speakerphone. This is an implementation choice, the selection of the desired microphone or the activating of the microphone could be tied to the selection of the desired audio device.

Fig. 2 shows details of the audio system within a multi-media workstation 7 according to Fig. 1. Workstation 7 comprises an input interface 71 which couples workstation 7 with network 1. Input interface 71 is coupled with a digital signal processor 7 which processes the packetized incoming data stream into a stereo audio signal and processes any audio signal from microphone 13 received through an A/D converter in audio card 73 into a packetized data stream which can be transferred over network 1. A program control unit 74, such as the central processing unit, receives data from a keyboard or mouse or any other input device and feeds a respective control signal to DSP 7. DSP 7 generates a left and a right audio signal which is fed to an audio card 73. Audio card 73 comprises D/A and A/D converter and respective amplifier stages to drive loudspeaker 8 and/or headset 10. The audio card may also have additional input/outputs 14, for example, for a second set of loudspeakers and/or additional microphones.

In the embodiment shown in Fig. 2, DSP 7 manages incoming and outgoing data streams under the control of program control unit 74. In this embodiment only one speaker 8 is used which is coupled to either the right or left output of audio card 73. The other output of audio card 73 is connected to headset 10. In case of an incoming call DSP 7 generates a ringing signal which is fed to the audio channel coupled with loudspeaker 8. In another embodiment the ringing signal can be applied to both loudspeaker 8 and headset 10 with appropriate volume settings for each of the outputs. On the screen of workstation 7 an Alert-Box will be generated as for example shown in Fig. 5. Such an Alert-Box indicates in addition to the ringing signal that an incoming call is waiting to be picked up. The user can, for example in a window-based graphical user interface, move an arrow 104 to select one of three choices. The selection is done by clicking on a respective button or icon 101, 102, or 103. Clicking on button 101 selects a speaker phone function. In this mode all incoming audio data are processed by DSP 7 into a single digital audio signal which will be converted into an analog audio signal by means of audio card 73. This single analog signal is then fed to speaker 8. Clicking on button 102 selects the headset mode. In this mode DSP 7

generates the audio signal on the other audio channel which will be converted into a single audio signal by means of audio card 73 and fed to headset 10. Clicking on button 103 cancels the incoming call.

Another embodiment could additionally allow for the sending of the audio outputs to both the speaker as well as the headset, for instance, for someone who is hard of hearing or needs higher volume.

Another embodiment of an audio subsystem according to the present invention is shown in Fig. 3. Again, a DSP 80 is provided which processes a digital data stream 88, for example packetized audio data. DSP 80 is under control of a control signal 87, for example, provided by a central processing unit of a workstation. DSP 80 generates a single audio signal which is converted into an analog signal by means of D/A converter 89. The analog signal is fed to a switch 81 which operates under control of a control signal 82. Switch 81 feeds the analog signal either to line 83 which connects to the input of amplifier stage 84 or to line 85 which connects to the input of amplifier stage 86. Output of amplifier stage 84 is coupled with loudspeaker 8 and output of amplifier stage 86 is coupled with headset 10.

In this embodiment only a single audio signal is generated. In case of an incoming call switch 81 couples the output of D/A-converter 89 with the input of amplifier stage 84 so that the ringing signal will be fed to the loudspeaker 8. According to the selection made by the user, as shown in Fig. 5, switch 81 couples either loudspeaker 8 or headset 10 with the respective audio signal.

Fig. 4 shows a third embodiment which can be implemented in existing systems, for example, as an automatic external switch box. In this case a standard audio I/O card 90 is used within a personal computer. Audio I/O card 90 generates a stereo analog output signal which is fed to terminals 92 and 93. External switch box 96 comprises two switches 94 and 95. Switches 94 and 95 are controlled by a control signal 91. Control signal 91 can be provided by audio I/O card 90, or by a parallel or a serial output port, or any other suitable port of the respective workstation. Switches 94 and 95 either

couple loudspeakers 8, 9 or headset 10 with terminals 92 and 93, respectively.

In an other embodiment one output could be taken to a recording device and the other to a speaker or headset. The selection of the recorder output made using the screen, keyboard, mouse, etc.

In all cases the system will revert to a known state at the end of the call, ready to process the next incoming call. Thus, the ringing signal cannot be missed by the user in case he does not wear the headset nor is there any danger of the user being exposed to excessive sound level, if he does not switchback prior to the next call. Implementation choices may also allow the user to switch between the speaker and headset modes at any time during the actual call. Upon selection, as described above in combination with Fig. 5, either the loudspeakers 8, 9 or the headset 10 is used as an output device.